## IN THE CLAIMS

Please amend the claims as follows:

1. (Currently Amended) A method of generating a monaural signal—(S) comprising a combination of at least two input audio channels—(L, R)signals, said method comprising the steps of:

dividing said at least two input audio signals into a plurality of sequential segments;

<u>summing</u>, for each of a <u>plurality of the</u> sequential segments (t(n))—of said audio <u>channels</u> (L,R) <u>signals</u>, <u>summing</u> (46) corresponding frequency components from respective frequency spectrum representations for each audio <u>channel</u> (L(k), R(k)) <u>signal</u> to <u>provide—form</u> a set of summed frequency components (S(k))—for each sequential segment;

calculating, for each of said plurality of the sequential segments, calculating (45) a correction factor (m(i)) for each of a plurality of frequency bands (i) as function of the energy of the frequency components of the summed signal frequency components in said band  $(\sum_{k \in I} |S(k)|^2)$  and the energy of said frequency components of

the input audio <code>channels\_signals\_in said band (  $\sum_{k \in l} \left| L(k) |^2 + |R(k)|^2 \right\})$  ; and</code>

correcting (47)—each summed frequency component as a function of the correction factor (m(i)) for the frequency band of said component; and

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outputting said corrected summed frequency components as said monaural signal.

 (Currently Amended) A—<u>The method according to as claimed in</u> claim 1, wherein said method further comprising—comprises the steps of:

providing (42)—a respective set of sampled signal values for each of a plurality of sequential segments for each input audio ehannelsignal; and

transforming, for each of said plurality of sequential segments, transforming (44) each of said set of sampled signal values into the frequency domain to provide said-complex frequency spectrum representations of each input audio channel (L(k),R(k))signal.

3. (Currently Amended) A—The method according to as claimed in claim 2, wherein the step of providing said sets of sampled signal values comprises:

<u>combining</u>, for each input audio <u>channelsignal</u>, <u>combining</u> overlapping segments (m1,m2)—into respective time-domain signals representing each <u>channel—input audio signal</u> for a time window (t(n)).

4. (Currently Amended) A—The method according—to as claimed in claim 1, wherein said method further comprising—comprises the step of:

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<u>converting</u>, for each sequential segment, converting (48) said corrected frequency spectrum representation of said summed signal (S'(k)) frequency components into the time domain.

5. (Currently Amended) A <u>The</u> method according to as claimed in claim 4, wherein said method further comprising comprises the step of:

applying overlap-add <del>(50)</del>—to successive converted summed signal representations to provide a final summed signal—<del>(s1,s2)</del>.

6. (Currently Amended)

A—The method according to as claimed in claim 1 wherein two input audio channels—signals are summed, and wherein said correction factors (m(i)) are determined according to the function:

$$m^{2}(i) = \frac{\sum_{k \in I} \left\{ L(k) \right|^{2} + |R(k)|^{2} \right\}}{2\sum_{k \in I} |S(k)|^{2}} = \frac{\sum_{k \in I} \left\{ L(k) \right|^{2} + |R(k)|^{2} \right\}}{2\sum_{k \in I} |L(k) + R(k)|^{2}} - \dots$$

7. (Currently Amended) A The method according to as claimed in claim 1, wherein two or more input audio channels ignals  $(X_n)$  are summed according to the function:

$$S(k) = C(k) \sum_{n} w_n(k) X_n(k)$$

wherein C(k) is the correction factor for each frequency component, and wherein said correction factors (m(i))—for each frequency band are determined according to the function:

$$m^{2}(i) = \frac{\sum_{n} \sum_{k \in i} |w_{n}(k)X_{n}(k)|^{2}}{n \sum_{k \in i} \left| \sum_{n} w_{n}(k)X_{n}(k) \right|^{2}}$$

wherein wn(k) comprises a frequency-dependent weighting factor for 10 | each input <u>channelaudio signal</u>.

- 8. (Currently Amended) A-The method according to as claimed in claim 7, wherein  $w_n(k)=1$  for all input audio channelssignals.
- 9. (Currently Amended) A-The method according to as claimed in claim 7\_ wherein  $w_n(k) \neq 1$  for at least some of the input audio channelssignals.
- 10. (Currently Amended) A—The method according to as claimed in claim  $7_{\perp}$  wherein the correction factor for each frequency component (C(k))—is derived from a linear interpolation of the correction factors (m(i))—for at least one band.
- 11. (Currently Amended) A—The method according to as claimed in claim 1, wherein said method further comprising—comprises the steps of:

 $\frac{\text{determining, }}{\text{bands, }} \text{ determining-an indicator } \left(\alpha(i)\right) - \text{of the phase difference}$  between frequency components of said audio  $\frac{\text{channels-signals}}{\text{channels-signals}}$  in a sequential segment; and

prior to summing corresponding frequency components,

transforming the frequency components of at least one of said audio

channels—signals as a function of said indicator for the frequency
band of said frequency components.

12. (Currently Amended) A-The method according to as claimed in claim 11, wherein said transforming step comprises operating the following functions on frequency components (L(k), R(k))—of left and right input audio ehannels (L,R) signals:

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$$L'(k) = e^{jca(i)}L(k)$$
$$R'(k) = e^{-j(1-c)a(i)}R(k)$$

wherein  $0 \cdot c \cdot 1$  determines the distribution of phase alignment between the said input ehannelsaudio signals.

- 13. (Currently Amended) A—The method according to as claimed in claim 1, wherein said correction factor is a function of a sum of energy of the frequency components of the summed signal in said band and a sum of the energy of said frequency components of the input audio channele—signals in said band.
  - 14. (Currently Amended) A component (S8')An apparatus for generating a monaural signal from a combination of at least two input audio channelssignals (L, R), comprising:

a segmenter for dividing said at least two input audio signals into a plurality of sequential segments;

a summer (46) arranged to sumfor summing, for each of a plurality of the sequential segments (t(n)) of said audio channels (t(n)) of said audio channels (t(n)) signals, corresponding frequency components from respective frequency spectrum representations for each audio channel (t(n)) signal to provide form a set of summed frequency components (t(n)) for each sequential segment;

means for calculating (45)—a correction factor (m(i))—for each of a plurality of frequency bands (i) of each of said plurality of sequential segments as function of the energy of the frequency components of the summed signal—frequency components in said band  $(\sum_{k \in I} |S(k)|^2)$  and the energy of said frequency components of

the input audio <code>channels\_signals\_in said band (  $\sum_{k \in I} \left\{ L(k) |^2 + |R(k)|^2 \right\})$  ; and</code>

a correction filter (47)—for correcting each summed frequency component as a function of the correction factor (m(i)) for the frequency band of said component, said correction filter outputting the monaural signal.

- 15. (Currently Amended) An audio coder including the <del>component</del> of apparatus as claimed in claim 14.
- 16. (Currently Amended) Audio An audio system comprising an audio coder as claimed in claim 15, and a compatible audio player.

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